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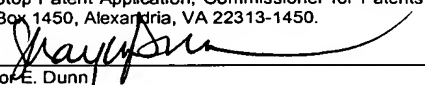
METHODS AND SYSTEMS FOR DETECTING AND MITIGATING CALL
ROUTING ARBITRAGE IN A TELECOMMUNICATIONS NETWORK

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Description

METHODS AND SYSTEMS FOR DETECTING AND MITIGATING CALL ROUTING ARBITRAGE IN A TELECOMMUNICATIONS NETWORK

5

Technical Field

The present invention relates to methods and systems for detecting and mitigating call routing arbitrage in a telecommunications network. More particularly, the present invention relates to methods and systems for detecting call signaling messages that have been routed and, in some cases, modified so
10 as to cause a call to be routed in an erroneous or fraudulent manner.

Background Art

In today's highly regulated telecommunications environment, network usage billing is governed by a complex set of rules and local access and
15 transport area (LATA) tariffs. In order to properly abide by these rules and tariffs, calls are normally routed over trunk groups that have been designated and configured to carry the particular call types. Examples of calls that have different tariffs include local calls, interLATA-interstate calls, interLATA-intrastate calls, intraLATA-interstate calls, intraLATA-intrastate calls, etc. In practice, this
20 complex tariff structure coupled with a large number of service providers and an associated array of complex interconnection agreements provides ample

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opportunity and incentive for calls to be inappropriately routed to their destinations. While such inappropriately routed calls may be the result of unintentional, erroneous routing translations in the network, all too often, such call routing is the result of a service provider trying to take advantage of the
5 complex tariff structure by exploiting the call routing limitations of current telecommunication networks.

For example, switches in the public switched telephone network (PSTN) route calls based solely on the called party number that is contained within the call setup messages exchanged between switches. In other words, network
10 switching elements make no attempt at determining the call type (e.g., local, interLATA-interstate, etc.). Hence, it is entirely possible for a network operator, intentionally or otherwise, to route any call type over any inter-switch trunk group.

It is possible, for instance, that calls which are non-local in nature may be routed over local trunk groups, thereby avoiding LATA access charges. An
15 example of such a call scenario is depicted in Figure 1. In Figure 1, an exemplary telecommunication network **100** includes a core PSTN network **102**, a tandem switching office **104** owned by Service Provider B, an end office **106** owned by Service Provider A, and a tandem switching office **108** owned by Service Provider C. An access trunk group **110** connects tandem switching
20 office **104** and end office **106**, while a local trunk group **112** connects tandem offices **104** and **108**, and a local trunk group **114** connects tandem office **108** and end office **106**. Network **100** also includes a calling party **116**, and a called party **118**, where the called party **118** is served by end office **106**.

For purposes of illustration, it is assumed that a call placed from calling party **116** to called party **118** is not a local call, and consequently should be subject to inter-LATA tariff charges. Accordingly, a call from calling party **116** to called party **118** should be routed from Service Provider B's tandem office **104** to the terminating end office **106** of Service Provider A via the access bearer trunk group **110** that has been designated to carry non-local call traffic. However, as indicated in Figure 1, Service Provider B may intentionally or otherwise route the non-local call to called party **118** via tandem office **108** over local bearer trunk groups **112** and **114**. As such, a non-local call may be completed between calling party **116** and called party **118** without using the access trunk(s) that have been designated to handle such non-local traffic by Service Provider A.

In cases where such misrouting is performed intentionally for financial gain, the service provider perpetrating the fraud may modify one or more fields in a call setup signaling message, such as a signaling system 7 (SS7) ISDN user part (ISUP) initial address message (IAM), so as to make the fraudulent routing difficult to detect. Figure 2 illustrates an ISUP IAM message **130**, which includes a calling party number (CgPN) parameter **132**, a called party number (CdPN) parameter **134**, and a jurisdiction information parameter (JIP) **136**, and a carrier identification code (CIDC) parameter **138**. One or more of these parameters may be modified in order to misroute a call and to hide the fact that fraudulent routing has occurred. For example, the JIP and/or calling party number parameters may be hidden or erased to prevent source identification for source-based billing.

Figure 3 is another view of network **100** that includes both bearer and signaling trunks. More particularly, signaling trunk **120** connects switches **104** and **108**, signaling trunk **122** connects switches **106** and **108**, and signaling trunk **124** connects switches **104** and **106**. In an SS7 network, IAM call setup messages are used by switches to select bearer trunks (e.g., bearer trunks **110**, **112**, **114**) and determine how a call will be routed within the network. In the sample call scenario depicted in Figures 1 and 2, if Service Provider B intends to route a call via local trunks **112** and **114**, IAM signaling messages are communicated from Service Provider B's tandem office **104** to Service Provider C's tandem office **108** via signaling trunk **120** and eventually to the terminating end office **106** via signaling trunk **122**. As described above, inter-carrier billing is often based on call origination information (e.g., calling party number, JIP parameter, etc.) Consequently, before misdirecting a call setup IAM message (as shown in Figure 2) and enabling the call to be completed without using the appropriate access trunk group **110**, Service Provider B's tandem office **104** may erase or alter a JIP or calling party number parameter contained in the call setup message. These parameters may be modified by Service Provider B in such a manner that it is not possible to determine the origin of the call, and, consequently, it is not possible for Service Provider A to determine that misrouting has occurred and/or the type of tariff to apply to the call.

The financial implications of such fraudulent routing may be substantial to a network operator. A typical routing arrangement associated with the network environment illustrated in Figure 1 might require Service Provider B to pay

Service Provider A terminating access charges in the range of \$.01 to \$.03 per minute for the call from calling party **116** to called party **118**. Whereas, in the case where an inter-LATA call is routed over local trunk groups, Service Provider B may only be required to pay a fraction of this per minute charge to Service
5 Provider A, if anything at all. The case where a call is intentionally misrouted to avoid tariffs is referred to as "arbitrage." However, the term "arbitrage," as used herein, is intended to include both the intentional and non-intentional misrouting of calls.

Accordingly, there exists a long felt need for improved methods and
10 systems for detecting, mitigating and preventing fraudulent or otherwise misrouted calls in a telecommunications network.

Disclosure of Invention

The present invention includes methods and systems for detecting,
15 mitigating and preventing the intentional or unintentional misrouting of calls in a telecommunications network. In one exemplary implementation, the present invention includes a network routing node, such as a signaling system 7 (SS7) signal transfer point (STP) or an Internet protocol (IP) capable signaling gateway (SG), which is adapted to intercept and examine call setup signaling messages.
20 More particularly, an arbitrage detector/processor associated with the routing node may examine parameters contained in an intercepted message and determine whether the message has been misrouted. If examined parameters in the message have been modified, erased, or otherwise contain insufficient

information to determine whether the message has been misrouted, the arbitrage detector/processor may reconstruct the modified message by obtaining the missing or modified information. The arbitrage detector/processor may also perform an arbitrage mitigation action. Arbitrage mitigation actions may include

5 redirecting the call to an interactive voice response (IVR) node collects the required information from the calling party. Once the necessary information is obtained and appropriate call routing can be confirmed, the call is completed to the called party. If a call is determined to be inappropriately routed, the IVR may inform the calling party of the problem and/or drop the call. The arbitrage

10 detector/processor may also generate a usage measurements and billing report that documents inappropriately routed and/or suspiciously routed call statistics.

An example of arbitrage that the present invention is intended to mitigate may include an IXC sending interstate traffic wants a connection to a terminating LEC but doesn't want to pay that LEC's tariff rates. The IXC can negotiate a

15 contract with an RBOC that allows the IXC to direct all its traffic to the RBOC, which then sends it on to the LEC. The RBOC and the LEC have a bill-and-keep arrangement rather than an access fee. So the IXC pays the RBOC a fee but avoids the higher tariff set by the LEC. Meanwhile, the RBOC gets to make a bundle from handling traffic it might not otherwise get. The RBOC may hide data

20 to prevent the LEC from identifying that the bypass is occurring. For example, the RBOC may change parameter values in signaling messages to masquerade long distance as local traffic to take advantage of bill-and-keep arrangements. The agreement between the RBOC and the long distance company is a formal

agreement; there's nothing hidden about it, except which calls are coming from the IXC. In order to avoid this type of arbitrage, the LEC has to identify incoming traffic. However, if the RBOC strips out the calling ID number (also known as the ANI) and other pertinent IXC information such as the CIC and jurisdictional
5 information, without the present invention, arbitrage cannot be identified.

Accordingly, it is an object of the invention to provide methods and systems for detecting and mitigating fraudulently or otherwise misrouted calls in a communications network.

It is another object of the invention to provide methods and systems for
10 reconstructing a call signaling message that has been altered in an attempt to fraudulently route a call.

It is another object of the invention to provide methods and systems for generating and reporting information associated with suspiciously routed or misrouted calls in a communications network.

15 Some of the objects of the invention having been stated hereinabove, and which are addressed in whole or in part by the present invention, other objects will become evident as the description proceeds when taken in connection with the accompanying drawings as best described hereinbelow.

20 Brief Description of the Drawings

Preferred embodiments of the invention will now be described with reference to the accompanying drawings of which:

Figure 1 is a network diagram illustrating an arbitrage call routing scenario in a telecommunications network;

Figure 2 is a block diagram illustrating a signaling system 7 ISDN user part (ISUP) Initial Address Message (IAM);

5 Figure 3 is a network diagram illustrating a message flow associated with an arbitrage call routing scenario in a telecommunications network;

Figure 4 is block diagram illustrating an exemplary architecture of a signaling gateway (SG) routing node suitable for use with embodiments of the present invention;

10 Figure 5 is a system block diagram illustrating a routing node including an integrated database services module for performing arbitrage detection and mitigation processing according to an embodiment of the present invention;

Figure 6 is a flow chart illustrating exemplary steps that may be performed in identifying and mitigating call arbitrage according to an embodiment of the
15 present invention;

Figure 7 is a functional block diagram of an arbitrage detector/processor according to an embodiment of the present invention;

Figure 8 is a network diagram illustrating messaging associated with a call release arbitrage mitigation action of the present invention;

20 Figure 9 is a network diagram illustrating call setup to an Interactive Voice Response (IVR) node associated with a call redirect arbitrage mitigation action of the present invention;

Figure 10 is a network diagram illustrating call setup in which a call for which arbitrage is detected is redirected to an IVR node associated with the called party end office according to an embodiment of the present invention;

Figure 11 is a block diagram and associated message flow of a routing
5 node including an integrated database services module for providing a number portability database according to an embodiment of the present invention;

Figure 12 is a network diagram illustrating number portability database access to determine call jurisdiction associated with one aspect of the present invention;

10 Figure 13 is a network diagram illustrating a media gateway controller including an arbitrage detector and an internal number portability database according to an embodiment of the present invention; and

Figure 14 is a network diagram illustrating an arbitrage detector and a number portability database associated with a network monitoring system
15 according to an embodiment of the present invention.

Detailed Description of the Invention

Disclosed herein are several embodiments of the present invention, some of which may include an underlying hardware platform similar to that of a signal
20 transfer point (STP) or an SS7-over-Internet protocol signaling gateway. Figure 4 is a block diagram illustrating an exemplary SG node **200**, which employs a distributed, multi-processor system architecture suitable for use with embodiments of the present invention. As shown in Figure 4, SG **200** includes

the following subsystems: a maintenance and administration subsystem (MAS) **202**, a communication subsystem **204** and an application subsystem **206**. MAS **202** provides maintenance communications, initial program loading, peripheral services, alarm processing and system disks. Communication subsystem **204** includes an interprocessor message transport (IMT) bus. The IMT bus facilitates communication among the various modules and subsystems that comprise the SG. In one exemplary implementation, the IMT bus includes two 1 Gbps counter-rotating serial rings. In an alternate implementation, communication subsystem **204** may include a local area network, such as an Ethernet, for connecting the modules and subsystems comprising SG **200**.

Application subsystem **206** includes application cards or printed circuit boards capable of communicating with the other cards through the IMT communications bus or network. Numerous types of application cards can be included in SG **200**. Exemplary application cards include an SS7 link interface module (LIM) **208** that provides SS7 links and X.25 links, a data communication module (DCM) **210** that provides an Internet protocol (IP) signaling interface to external nodes, and a high-speed asynchronous transfer mode (ATM) communication link module (HSL) **212**. A database services module (DSM) **214** may host one or more signaling message processing applications, such as global title translation, flexible routing, number portability translation, and an arbitrage detector according to the present invention.

The architecture illustrated in Figure 4 may be used to implement an arbitrage screening and mitigation application of the present invention for ISDN

user part (ISUP) messages as well as IP-encapsulated or IP-adapted ISUP signaling messages, such as Internet Engineering Task Force (IETF) SIGTRAN M3UA protocol messages. An arbitrage screening and mitigation application of the present invention may also process other non-SS7 signaling protocol
5 messages, such as session initiation protocol (SIP) signaling messages, media gateway control protocol (MGCP) signaling messages, MEGACO signaling messages, H.323 signaling messages, and the like.

Figure 5 illustrates an SG routing node **300** including an arbitrage detector/processor located on a database services module according to an
10 embodiment of the present invention. In Figure 5, SG routing node **300** includes a high speed IMT communications bus **302** and a pair of MASP processor modules **304**. MASP pair **304** implement the maintenance and administration subsystem functions described above. As MASP pair **304** are not essential to a discussion of the arbitrage function of the present invention, a detailed discussion
15 of their operation is not provided herein. IMT bus **302** may be a counter-rotating dual ring bus or an Ethernet, as described above.

A number of distributed processing communication and application modules or cards may be coupled to IMT bus **302**. In Figure 5, these processing modules or cards include an IP-capable DCM **306**, an ATM-capable HSL
20 communication module **308**, a pair of SS7 LIMs **310** and **312**, and a plurality of DSMs **314**, **316**, and **318**. DSM **314** may include an arbitrage detector/processor **320** according to an embodiment of the present invention. DSM **316** may include

a number portability translation database **321** for storing jurisdictional information useful in mitigating call arbitrage.

Each processing module in Figure 5 may include hardware, software, and/or firmware for performing the indicated functions. For example, in one
5 implementation, each module may include an application processor for processing and routing signaling message and a communications processor for inter-processor communication via bus **310**. Providing a distributed processing architecture, such as that illustrated in Figure 5, increases message throughput and decreases on-the-fly processing time for each message, even when
10 arbitrage screening and mitigation is performed.

In Figure 5, a single DSM is provisioned to provide arbitrage detection processing functionality within SG **300**. However, multiple DSMs with arbitrage detection may be simultaneously deployed within SG **300** without departing from the scope of the invention. In addition, arbitrage detector/processor **320** may be
15 located on the link interface modules, such as LIM **310** or DCM **306**, without departing from the scope of the invention.

As stated above, LIMs **310** and **312** communicate with other nodes via external SS7 signaling links. As such, each LIM may be connected to other signaling points in a network via one or more individual SS7 signaling links,
20 where an SS7 signaling link is typically a 56 kbps or 64 kbps DS-0 link. Multiple signaling links connected to a common destination may be grouped into a virtual entity known as an SS7 signaling linkset.

DCM **306** includes IP transport capabilities. DCM **306** may be connected to external nodes via one or more IP-based signaling linksets. DCM **306** may transmit any type of signaling messages over the IP based signaling linksets. Examples of signaling messages that may be transmitted over IP signaling links by DCM **306** include SS7 signaling messages, SIP signaling messages, and H.323 signaling messages.

As illustrated in Figure 3, LIM **310** includes an SS7 MTP level 1 & 2 function **322**, a message discrimination function **324**, a routing function **326** and a message distribution function **328**. MTP level 1 and 2 function **322** provides the facilities necessary to send and receive digital data over a particular physical medium, as well as to provide error detection, error correction and sequenced delivery of SS7 messages. Message discrimination function **324** receives signaling messages from the lower processing layers and performs a discrimination function that determines whether an incoming SS7 message requires internal processing or is simply to be through switched. In one exemplary implementation, discrimination function **324** may perform an arbitrage pre-screening function to identify messages as candidates for arbitrage processing. For example, discrimination function **324** may examine the SI and ISUP message type parameters to identify ISUP IAM messages as candidates for arbitrage processing. In an alternate implementation, discrimination function **324** may determine whether a signaling message is a candidate for arbitrage processing based on the signaling link or signaling linkset on which the message is received. The remaining ISUP originating party and jurisdictional information

parameters may be examined by discrimination function **324** and/or arbitration detector/processor **320** to determine whether the call associated with the message is being improperly routed. Performing arbitration prescreening on the LIMs is advantageous because such prescreening distributes arbitration detection
5 processing over multiple processors.

DCM **306** may also include a discrimination function that identifies IP-based signaling messages as candidates for arbitration processing. For IP-based SS7 signaling messages, the discrimination function on DCM **306** may examine the same parameters that discrimination function **324** on LIM **310** examines. For
10 other types of signaling messages, DCM **306** may examine parameters that determine whether the message type is an initial call signaling message and therefore a candidate for arbitration processing. For example, for SIP messages, the discrimination function on DCM **306** may identify INVITE messages as candidates for arbitration processing. For H.225 messages, the discrimination
15 function on DCM **306** may identify CONNECT messages as candidates for arbitration processing. Subsequent arbitration screening for the identified candidates may be performed by the discrimination function on DCM **306** or by arbitration detector/processor **320**, as will be described in detail below.

For received signaling messages that simply require MTP routing, routing
20 function **326** is responsible for examining these messages and determining on which outbound MTP signaling link or IP signaling link over which these messages are to be transmitted. Routing function **326** may transmit the

messages to the outbound communication module (e.g., LIM, DCM, HSL) associated with the selected signaling link via IMT bus **302**.

If discrimination function **324** determines that a received signaling message requires processing by an internal application processor or associated subsystem of SG node **300**, such as an arbitration application, then the message is passed to message distribution function **328**. An associated subsystem may be an adjunct message processing platform, such as a computer workstation or server, that is connected to the SG via an external communication connection, such as an Ethernet connection. For example, discrimination function **324** may determine that a received SS7 signaling message requires arbitration processing if the message includes an SI parameter value of 5, indicating that the message is an ISUP message. In one implementation discrimination function **324** may decode the ISUP parameters in the signaling message to produce a decode key and forward the decode key along with the signaling message to arbitration detection DSM **314** via IMT bus **302**. Arbitration processing may be performed by DSM **314**, which may result in some or no modification of a received signaling message and/or, in some cases, the creation of a new signaling message. Upon completion of arbitration processing, messages may be directed from DSM **322** to an outbound LIM via IMT bus **302**, as generally indicated in Figure 5.

The present invention is not limited to the specific architecture illustrated in Figure 5. Multiple LIMs, DCMs, HSLs, DSMs, and other types of communication and processing modules may be included without departing from the scope of the invention.

Figure 6 is a flow chart illustrating exemplary steps that may be performed in identifying and mitigating call arbitrage according to an embodiment of the present invention. The steps illustrated in Figure 6 may be performed by arbitrage detector/processor **320** and/or discrimination function **324** illustrated in Figure 5. The particular distribution of processing responsibility between these functions may depend on the relative processing load between LIM **310** and DSM **314**. In one exemplary implementation, discrimination function **324** may perform initial arbitrage pre-screening and arbitrage detector/processor **320** on DSM **314** may perform the arbitrage screening and mitigation operations.

Referring to Figure 6, in step **330**, a plurality of signaling messages are received and candidates for arbitrage screening are identified. Step **330** may be performed by examining the message type. If the message type indicates that the signaling message is an initial call setup signaling message, such as an IAM message, an INVITE message, a CONNECT message, etc., the signaling message may be identified as a candidate for arbitrage screening.

In step **332**, for the signaling messages identified as candidates for arbitrage screening, arbitrage screening is performed. This step may include examining one or more originating party parameters and/or jurisdictional information parameters in the candidate signaling messages to determine whether the parameters have been altered or have been deleted from the signaling messages. As discussed above, one method for performing call arbitrage is to replace or nullify calling party parameters in ISUP IAM messages. Thus, if a received ISUP IAM message does not include a calling party number,

call arbitration may be identified. In another example, determining whether a calling party number is valid may include comparing the calling party number to a list or range of expected or valid calling party numbers. If the received calling party number is not within the range or list, the calling party number is identified
5 an invalid and arbitration is indicated. In another example, identifying the presence of call routing arbitration may include comparing the calling party number to a value or range of values known to be used for call routing arbitration. The values that arbitration detector/processor 320 uses to detect the presence of call routing arbitration are preferably customizable by the network operator.
10 Similar detection rules may be used to check other parameters in a signaling message, such as JIP or CIC parameter.

In step 334, if call arbitration is not identified, control returns to step 330 where the pre-screening of signaling messages continues. However, if call arbitration is identified in step 334, control proceeds to step 336 where a mitigating
15 action is performed. The particular mitigating action performed depends on the network implementation. Exemplary mitigating actions that will be described in detail in the message routing scenarios described below include releasing calls, redirecting calls, and reconstructing signaling messages to complete calls for which call arbitration has been identified.

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Arbitration Detector/Processor

Figure 6 is a functional block diagram of an arbitration detector/processor according to an embodiment of the present invention. In the illustrated example,

arbitrage detector/processor **320** includes a screening function **340**, a call release function **342**, a call redirect function **344**, a call pass function **346**, and a message accounting and reporting function **348**. Screening function **340** may examine calling party number and JIP parameters contained in a received ISUP
5 IAM signaling message, and determine whether that these parameters contain null or invalid values. Screening function **340** may also examine other information that is located in or implied by a received signaling message. Examples of other parameters located in an ISUP IAM message that may be examined include a carrier identification code parameter and an originating node
10 address, such as an originating point code. Examples of information implied by a received signaling message include the signaling link or linkset on which a message is received. For received SIP messages, screening function **340** may examine the SIP from parameter to determine whether the message contains a valid originating SIP URI. Examining any type of originating party information in
15 a received signaling message to identify the presence of arbitrage is intended to be within the scope of the invention.

If screening function **340** determines that arbitrage has occurred, screening function **340** may direct a screened signaling message to one of the provisioned functions **342**, **344**, or **346**. The particular function depends on the
20 network implementation and is preferably configurable by the network operator. For example, some network operators may desire to release all calls for which arbitrage is detected. Accordingly, in such networks, the signaling messages associated with these calls would be directed to release function **342**. Other

network operators may desire to reconstruct call signaling messages and route calls using the reconstructed call signaling messages. In this situation, screening function **340** may redirect signaling messages for which arbitrage has been detected to redirect function **344**. Finally, other operators may desire to pass

5 calls for which arbitrage has been detected and generate accounting records for later analysis. In such situations, screening function may redirect signaling messages for which arbitrage has been detected to pass function **346**. A single network operator may configure different arbitrage mitigation actions to be performed depending on the type or severity of call routing arbitrage. For

10 example, a network operator may initially configure arbitrage detector/processor **320** to pass all calls for which arbitrage has been detected but to generate reports. Upon analysis of the reports, the network operator may identify a particular carrier as a repeat offender. In this situation, the network operator may re-configure arbitrage detector/processor **320** to block calls from the repeat

15 offender and to pass other calls, while generating reports.

As indicated above, call release function **342** may prevent the setup of a call identified as being subject to arbitrage. For example, call release function **342** may simply discard a received call setup signaling message, thereby terminating call setup. In an alternate implementation, screening function **340**

20 may forward the signaling message to release function **342**, and release function **342** may generate and return a release message so as to more gracefully terminate an arbitrage-related call setup attempt. In an SS7 signaling environment, release function **342** may generate an ISUP REL message that is

associated with a received ISUP IAM message. Release function **342** may include within the REL message a release cause code, which indicates that call setup is being terminated because arbitrage is suspected or because insufficient information (e.g., CgPN, JIP, etc.) is present in the received ISUP IAM message.

- 5 Release function **342** may also notify message accounting and reporting function **348** of such call setup termination activity, whereby function **348** is adapted to record and report information associated with the release mediation action. Information recorded and reported by function **348** may include some or all of the received call setup message, the receipt time, the receipt date, and receiving or
10 inbound link or linkset. Peg count type information associated with arbitrage related call setup messages may also be recorded and reported by function **348**.

For non-SS7 calls, release function **342** may either discard received signaling messages for which arbitrage has been detected or send a release message in the appropriate protocol type. For example, for SIP signaling,
15 release function **314** may send a BYE or CANCEL message to end the call setup. In H.225 signaling environment, release function **342** may send a release message in response to signaling messages for which arbitrage has been detected.

Call redirect function **344** may redirect a call that suspected of arbitrage
20 routing in such a manner that the calling party is connected to an interactive voice response (IVR) server or a tone and announcement server (TAS). In the case of redirection to a TAS, the serving TAS may play the calling party a message, which explains why the call cannot be completed. In the case of

redirection to an IVR, the serving IVR may request and collect information from the calling party prior to completion of the call to the original called party. If the requested information is successfully collected by the IVR server, call setup to the original called party may continue. If the requested information is not

5 collected by the IVR server, the call may be terminated. Exemplary information collected from a calling party may include, the calling party telephone number, calling party geographic location information, and calling party service provider and/or carrier identification information. In a manner similar to that described above with respect to release function **342**, redirect function **344** may notify

10 message accounting and reporting function **348** of such call redirection activity. Message accounting and reporting function **348** may record and report information associated with the redirect action. Information recorded and reported by function **348** may include some or all of the received call setup message, the receipt time, the receipt date, the receiving or inbound link and/or

15 linkset, the redirected IVR identifier, and/or the redirected TAS identifier. The redirect IVR or TAS nodes may also record and report similar information associated with the processing of arbitrage related calls.

Call pass function **346** may notify message accounting and reporting function **348** of a call that suspected of arbitrage routing and subsequently route

20 the call to the specified called party. In this case, a call setup attempt that is believed to be associated with an arbitrage scenario is simply noted or recorded by arbitrage detector **320** for future reporting or analysis.

Exemplary Arbitrage Call Release Scenario

Figure 8 is a message flow diagram illustrating an example of call release arbitrage scenario that may be implemented by arbitrage detector/processor **320**. In Figure 8, network **400** includes an arbitrage-detector-equipped SG node **300**, similar to that illustrated in Figure 5. Network **400** also includes a switching office **402** associated with a first service provider, Service Provider A and a switching office **404** associated with a second service provider, Service Provider B. The illustrated signaling message flow includes a first ISUP IAM message **406**, which is received at tandem switching office **404**. ISUP message **406** includes a JIP parameter value of 919380, and a calling party number parameter value of 9194605500. In response to receiving ISUP message **406**, tandem switching office **404** generates a new, related, ISUP IAM signaling message **408**. Signaling message **408** is formulated by switch **404** such that the JIP and calling party number parameters are erased and set to null values. Message **408**, which is intended to establish a voice circuit between office **404** and office **402**, is transmitted to SG **300** for subsequent routing to office **402**. However, upon receipt of message **408**, SG **300** screens the message and determines that the message requires arbitrage processing, detects the null VIP and calling party parameters, discards the message, and sends a release message **410** indicating that the call has been released because of arbitrage.

Referring to the system diagram of SG **300** in Figure 5, message **408** is received at SG **300** by LIM **310**. Lower level MTP processing is performed via MTP level 1 and 2 function **322**, and the message is passed to discrimination

function **324**, where arbitration pre-screening is performed. Since the message is an IAM message, discrimination function **310** forwards the message to arbitration detector/processor **320**. Arbitration detector/processor **320** examines the contents of message **408** to determine whether calling party number and JIP parameter information is present and valid. In this example, it is determined that CgPN and JIP parameter information is not present, and screening function **340** therefore directs the message to release function **342**. Release function **342** discards the IAM message and formulates a release message indicating that the reason the call was dropped was arbitration.

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Exemplary Arbitration Call Redirect Scenario

Figures 9 and 10 illustrate an example of call redirection in response to arbitration according to an embodiment of the present invention. In Figure 9, network **400** includes an arbitration detection and processing equipped SG node **300**, similar to that described above and illustrated in Figure 5. Network **400** also includes an IVR node **416** to which arbitration related calls are redirected. As in the previous example, the illustrated signaling message flow includes a first ISUP IAM message **406**, which is received at tandem switching office **404**. ISUP message **406** includes a JIP parameter value of 919380 and a calling party number parameter value of 9194605500. In response to receiving ISUP message **406**, tandem switching office **404** generates a new, related, ISUP IAM signaling message **408**. Signaling message **408** is formulated by switch **404** such that the JIP and calling party number parameters are erased and set to null

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values. Message **408**, which is intended to establish a voice circuit between office **404** and office **402**, is transmitted to SG **300** for subsequent routing to office **402**. However, upon receipt of message **408**, SG **300** screens the message and determines that the message requires arbitration processing.

5 The screened ISUP IAM message **408** is directed to arbitration detector/processor **320** for arbitration processing. In this case, ISUP IAM message **408** is directed by screening function **340** to call redirect function **344**. Call redirect function **344** moves the original called party telephone number information from a called party number parameter in the ISUP IAM message and
10 places the number in another location within the IAM message, such as a generic address parameter (GAP). Subsequently, a routing number associated with an IVR server (or TAS) is inserted in the called party number parameter of the IAM message, and the modified message **410** is routed from SG **300** to switch **402**. Switch **402** receives IAM message **410**, selects a voice trunk and routes an
15 associated IAM message **412** to IVR **416** via SG **300**. A voice connection is established between the calling party and arbitration IVR node **416**.

 Arbitration IVR node **416** requests and collects information associated with the call from the calling party. For example, IVR **416** may play a recorded message, which requests that the calling party to specify his or her phone
20 number either by speaking the number or by using a telephone keypad. The IVR may also request that the calling party identify the calling party's long-distance carrier. If this information is successfully collected, the information may be inserted into an associated ISUP IAM message, which is in turn used to complete

call setup to the original called party (i.e., the telephone number that was temporarily stored in the GAP parameter).

If IVR **416** successfully collects the dialed digits, tandem office **402** may formulate an IAM message **412** with the appropriate dialed digits in the message and forward IAM message **412** to called party end office **418** via signaling gateway **300**. Since the IAM message includes a valid calling party address, the message may pass arbitration screening and be forwarded to end office **418**. End office **418** may then complete the call using standard call setup procedures.

If IVR **416** is not successful in collecting the calling party dialed digits, IVR **416** may play an announcement to the calling party indicating that the call will not be completed and instruct tandem office **402** to drop the call. Alternatively, tandem office **402** may allow the call to be completed and arbitrage detector **320** may simply generate a report so that the service provider can identify that arbitrage has occurred and prevent future instances of call arbitrage without dropping any calls in progress.

Figure 10 illustrates an alternate implementation of the invention in which IVR **416** is located at called party end office **418**. In the message flow illustrated in Figure 10, service provider B **404** nullifies calling party information in an IAM message, as described in the previous examples. Arbitrage screening function **320** detects the presence of call arbitrage and routes the call to IVR **416** associated with end office **418**. IVR **416** attempts to collect the calling party dialed digits. If IVR **416** is successful in collecting the calling party dialed digits, the call may be completed as normal. On the other hand, if IVR **416** is not

successful in collecting the calling party dialed digits, IVR **416** may play an announcement indicating that the call will not be completed, and end office **418** may drop the call. Alternatively, end office **418** may complete the call, and arbitration detector **320** may generate a report to the service provider to prevent
5 future call routing arbitration. Thus, in the embodiment illustrated in Figure 10, where IVR **416** is located at the called party end office, the amount of processing performed by arbitration detector **320** is reduced because arbitration detector **320** is only required to process a single IAM message.

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Arbitration With Number Portability

Even with an IVR redirect, such as that described above, it is possible that collected calling party information may not be sufficient to completely mitigate or prevent arbitration call routing. The widespread implementation of number portability within telephone networks (both wireline and wireless) present such a
15 challenge. In cases where a calling party has been ported (e.g., geographically, between service providers, etc.), the calling party telephone number may be insufficient to identify and/or mitigate call routing arbitration. In these cases, a jurisdiction identification parameter (JIP) contained in an ISUP IAM message may be necessary to resolve questions of call routing arbitration. With this
20 problem in mind, a slight variation of the above described arbitration mitigation processes (i.e., call pass, call redirect) may be implemented in such a manner as to obtain or verify the JIP parameter value in a received ISUP signaling message.

In one embodiment, SG node **300** may access a number portability (NP) database during arbitrage processing. The NP database may be co-located with SG **300** or the NP database may reside elsewhere in the signaling network or a data network. In the example shown in Figure 11, arbitrage detector/process **320**

5 queries an internal NP database **321** and thereby obtain and/or verify a JIP parameter value associated with a call. The JIP value associated with a call is usually the first six digits (i.e., NPA-NXX) of the location routing number (LRN) that is assigned to the switch that originates a call. It is also possible that a JIP data field may be included in NP database **321** if the first six digits of the LRN are

10 not used as the JIP value. In any event, NP database **321**, which associates subscriber telephone numbers with switch LRN values (or JIP values) may be used to resolve a JIP parameter value if one is not present, or has been stripped from a signaling message. Table 1 shown below illustrates subscriber telephone number information and associated LRN data that may be contained in number

15 portability database **321**. Using the exemplary data shown in Table 1, a JIP parameter value of 919380 may be determined from an NP database lookup for a call associated with a calling party whose telephone number is 9194605500.

Subscriber Telephone Number	LRN
9194605500	919380 1000
9194605501	9194600000
9194605502	9194600000
9194605503	9194600000

Table 1: Sample Number Portability Data

Figure 12 illustrates a variation of the arbitration processing scenario described above with respect to Figure 9. In Figure 12, it is assumed that SG 300 has identified arbitration based on an IAM message, that SG 300 has forwarded the IAM message to IVR 416, and that IVR 416 has collected calling party telephone number information and inserted this information into ISUP IAM message 420, which is routed to SG 300. SG 300 intercepts message 420 and directs the message to arbitration detector/processor 320 for processing, as indicated in Figure 11. Arbitration detector/processor 320 queries NP database 321 and obtains the correct JIP parameter value of 919380. The JIP parameter value is inserted into the modified ISUP IAM message 424. The JIP parameter may also be recorded and reported by message accounting and reporting function 348 (illustrated in Figure 7). Modified IAM message 424 is routed from SG 300 to switch 404, which in turn generates a related IAM message 426, which is routed to the end office 418 that is serving called party 450. Determining the correct jurisdiction of a call enables arbitration call routing to be much more easily identified and mitigated.

In the example illustrated in Figure 11, arbitration detector/processor **320** queries NP database **321**. In an alternate embodiment, IVR node **416** may directly query an NP database in order to verify JIP parameter information following collection of calling party information. The NP database may be co-located with the arbitration mitigation equipped SG **300** or may reside elsewhere in the network.

In cases where the selected arbitration mitigation action is the “pass” function, an NP database query similar to that described above may be performed. If the received signaling message is missing a JIP parameter, the JIP information returned from the NP database query may be inserted into the received message prior to routing the message from the SG node. As such, the call setup attempt is not blocked by SG **300**, but the correct JIP information is inserted into the call setup message, and may be later used by Service Provider A to determine if arbitration has occurred. The JIP information returned from the NP database lookup may also be recorded and reported by message accounting and reporting function **348**.

Soft Switch/Media Gateway Controller Implementation

Figure 13 illustrates an alternate implementation of an arbitration detection and system of the present invention, which is deployed in a packet switched communications network environment. In Figure 13 a packet network **500** includes a media gateway controller (MGC) or soft switch **502**, a tandem switching office **402** owned by Service Provider B, an IVR node **416**, an end

switching office **418** owned by Service Provider A, a called party **450**, a first media gateway (MG) **504**, a second MG **506**, and a packet network **508**. MGC node **502** sends and receives control signals to MG nodes **504** and **506** using a media gateway control protocol, such as MGCP or MEGACO. MG node **504** and
5 **506** provide an interface between circuit-switched time division multiplexed (TDM) communications and packet switched communications in the network. MGC node **502** also supports SS7 signaling as well as other call setup signaling protocols (e.g., SIP, H.323, IETF SIGTRAN SS7-over-IP, etc.).

In Figure 13, an arbitration detector/processor **320** and a number portability
10 database **321** are co-located with or accessible by MGC **502** in a manner similar to that described above with respect to the SG routing node **300**. Arbitrage detector/processor **320** associated with MGC **502** includes a call release function, a call redirect function, a call pass function, and message accounting and reporting function as described above with regard to Figure 7. IVR node **416**
15 may be used in conjunction with MGC **502**, in a manner similar to that previously described with respect to SG routing node **300**, to request and collect calling party information prior to call setup completion. As described above, NP database **321** may be used to obtain jurisdiction information associated with an attempted call.

20 As the fundamental signaling message flows and message processing operations are nearly identical between the MGC embodiment and the SG routing node embodiment, a detailed discussion of these flows and operations is not repeated herein for the MGC embodiment. The present invention may be

practiced in a similar manner at both network routing nodes, such as signaling gateways or signal transfer points, and MGC or soft switch nodes.

External Monitoring System Embodiment

5 In an alternate implementation of the present invention, arbitrage detection and processing operations may be performed by a stand-alone signaling network monitoring system that generates call detail records (CDRs) and copies of call setup signaling messages to an arbitrage detection and processing application. Figure 14 illustrates an arbitrage detector/processor **320** configured to operate
10 with a stand-alone network monitoring system. In Figure 14, a signaling message monitoring system **600** copies signaling messages from signaling links **122** and **124**. Monitoring system **600** correlates copied signaling messages and produces CDRs for each observed call or call attempt. Arbitrage detector/processor **320** is coupled to network monitoring system **300** via a
15 communications link, such as an IP communications link. Arbitrage detector/processor **602** may also be coupled to a number portability database **321** to obtain jurisdictional information when that information is omitted from a signaling message, as described above.

 In operation, network monitoring system **600** copies signaling messages
20 and generates CDRs based on the signaling messages. Network monitoring system **600** may forward the CDRs and/or the signaling message copies to arbitrage detector/processor **320**. Arbitrage detector/processor **320** screens the signaling messages to identify calls where arbitrage has occurred. In response

to detecting that arbitrage has occurred, arbitrage detector/processor **320** may generate reports to identify to the service provider that arbitrage is occurring in their network or perform any of thither arbitrage mitigation actions described above. The system illustrated in Figure 14 is particularly advantageous because
5 it can be deployed in an existing network without modification to current network elements.

Thus, the present invention includes improved methods and systems for detecting call routing arbitrage and performing mitigating actions once the arbitrage is identified. For example, an arbitrage detector/processor according to
10 the present invention may analyze signaling messages to determine whether one or more parameters in the signaling messages indicate the presence of call routing arbitrage. In response to determining that one of the parameters in the signaling messages indicates the presence of call routing arbitrage, the arbitrage detector/processor **320** may perform an arbitrage mitigation action, such as
15 dropping the call notifying the calling party, or obtaining the missing originating party information and completing the call using the missing originating party information. Because arbitrage can be detected and a mitigating action can be performed by a single network entity, the present invention provides an effective way to reduce telecommunications service provider revenue loss due to call
20 routing arbitrage.

It will be understood that various details of the invention may be changed without departing from the scope of the invention. Furthermore, the foregoing

description is for the purpose of illustration only, and not for the purpose of limitation--the invention being defined by the claims.